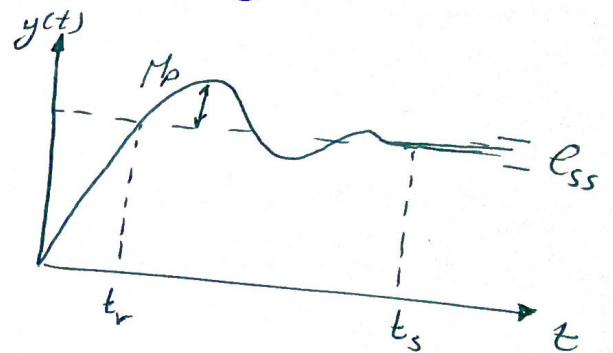


Digital Control (Session 1)

* Review (Control Engineering)

- Control theory is the general framework for studying dynamical systems
- Control Systems Objective is to improve the behaviour of the system to meet design specs (improve dynamics - reduce error)
- Any Control System Response Can be characterized by 4-parameters:

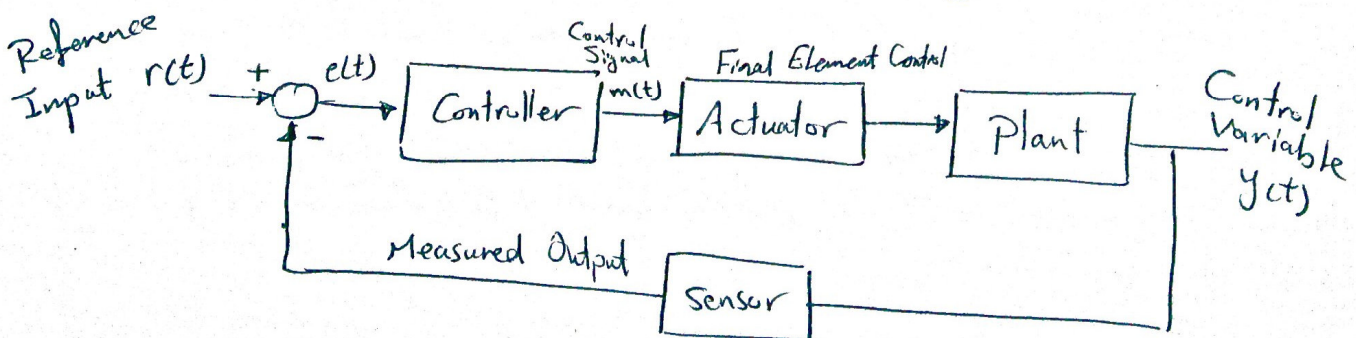
t_s : settling time
 t_r : rise time
 M_p : Maximum Overshoot
 e_{ss} : steady State Error



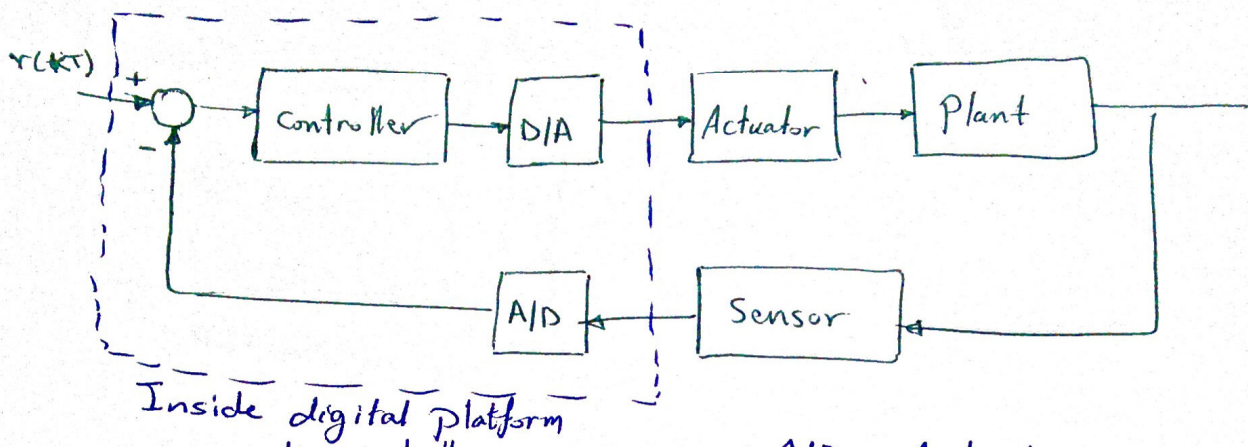
- The Phases of Control System design

- Plant Modeling (differential Equations - Transfer function)
- System Analysis (Root Locus - Frequency Response)
- Controller design (PID - Lead/Lag Compensator - State feedback)
- Implementation (Analog Components (op-amps + RLC circuits))

- The Block diagram of a Feedback Control System



* Digital Control System Structure



Inside digital platform

- Microcontroller
- DSP
- Computer
- PLC

A/D: Analog to Digital Converter

D/A: Digital to Analog Converter

* Digital platforms are the most powerful computing platforms specially with today's high speed of processors (much more than speed of systems to be controlled).

* Digital Control Benefits:

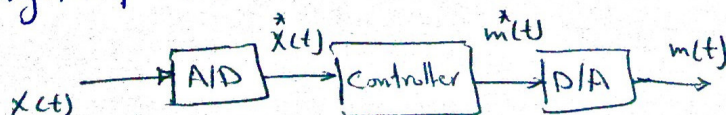
- Control Algorithm is converted to a code on a computer system
- More powerful to implement complex control algorithms (there are some control algorithms that can be implemented only using digital control techniques)
- Reduce the need of analog components (affected by noise).
- More efficient to be modified and scaled
- More robust to environment disturbance

* Digital Control challenges:

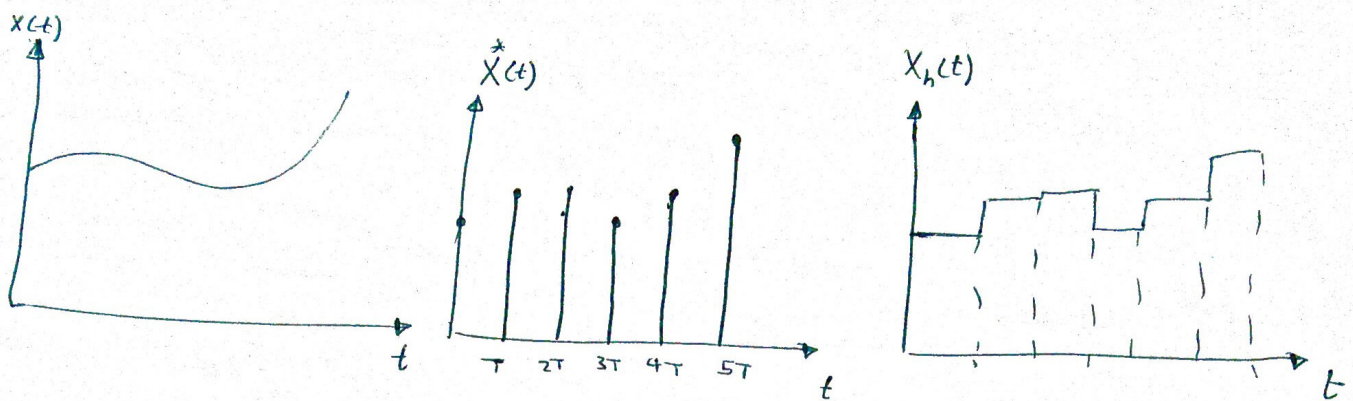
- Sampling Rate
- Reconstruction of original signals

Physical systems are continuous by nature we approximate them by discretization

* Digital part:



* Sampling and Reconstruction



$x(t)$: Original Analog Signal
 $x^*(t)$: Sampled Signal
 $x_h(t)$: Reconstructed Signal

$x_h(t)$: the reconstructed signal using
 Holder circuit is a ladder signal
 It can be smoothed using
 Low Pass Filter

We note that

$$\begin{aligned}
 x^*(t) &= \sum_{k=0}^{\infty} \delta(t - kT) x(kT) \quad , \quad T: \text{Sampling Period} \\
 &= \delta(t) x(0) + \delta(t - T) x(T) + \dots \\
 &= x(0) + x(T) + x(2T) + \dots
 \end{aligned}$$

Remember Transfer Functions describing Systems using Laplace Transform

$$\mathcal{L}[x^*(t)] = \sum_{k=0}^{\infty} x(kT) e^{-kTS} \quad (\text{Integration becomes Summation})$$

Note : This is the definition of Z-transform

$$Z = e^{kTS} \quad (Z: \text{advance operator}, Z^{-1}: \text{delay operator})$$

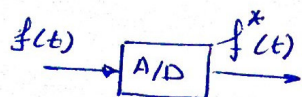
\therefore We need to make revision on Z-transform

- * Z-transform represents discrete Systems Transfer function
- * Difference equations are solved using Z-transform

* Sampling Process :-

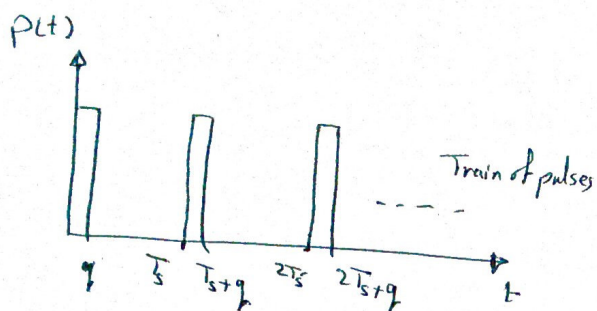
→ Is there a relation between Sampling frequency and Signal frequency?

For A/D Converter:



$$f(t) \xrightarrow{T_s} f^*(t) \quad (q \ll T_s)$$

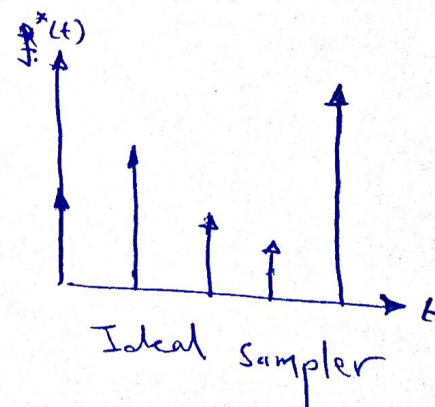
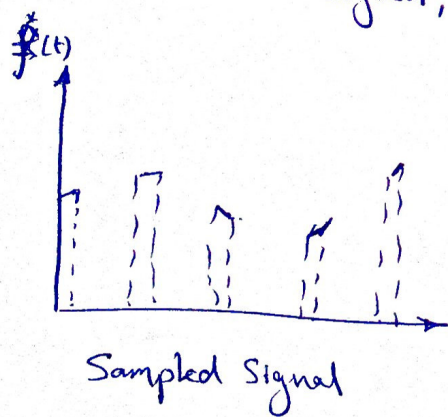
$(q=0 \text{ for Ideal Sampler})$



$p(t)$: models the opening/closing of Sampler Switch

then, $x^*(t)$ is the result of multiplying $x(t)$ by $p(t)$

* The Sampler is considered as an amplitude Modulation device with $p(t)$ as the Carrier Signal, $f(t)$ the input and $f^*(t)$ is the output signal.



$$f^*(t) = p(t) f(t)$$

$$(q \ll T_s)$$

where

$p(t)$: periodic function

$$p(t) = \sum_{k=-\infty}^{\infty} U(t - kT) - U(t - (kT + q))$$

q : t_{on}

T_s : Sampling Period

* The relation between f_s and f_o

f_s : Sampling frequency

f_o : Signal frequency

We use Fourier Analysis to derive the relation between f_s and f_o .

① Express $P(t)$ using Fourier Series ($P(t)$: periodic)

$$P(t) = \sum_{n=-\infty}^{\infty} C_n \exp(jn\omega_s t)$$

$$C_n = \frac{1}{T} \int_0^T P(t) \exp(-jn\omega_s t) dt$$

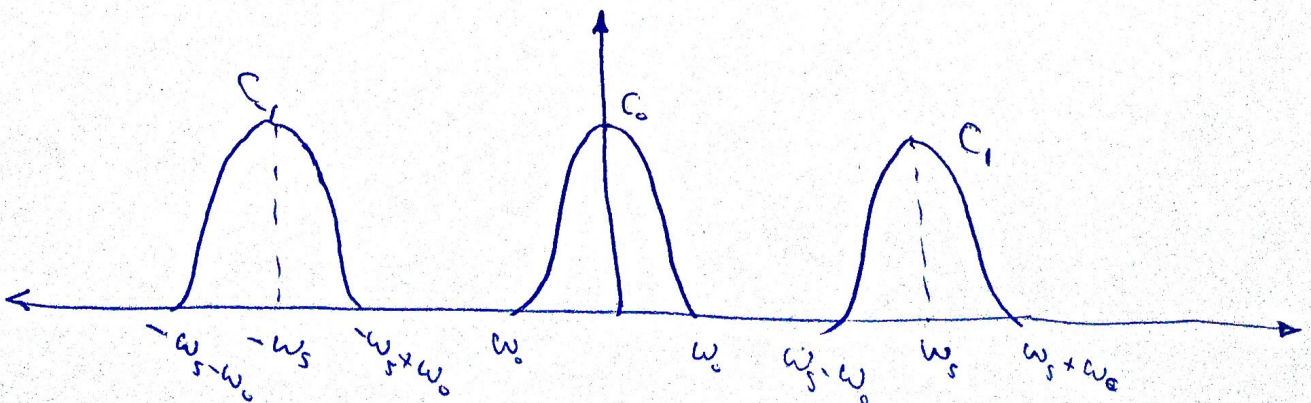
$$\therefore f^*(t) = f(t) \sum_{n=-\infty}^{\infty} C_n \exp(jn\omega_s t)$$

② Taking Fourier transform of $f^*(t)$ (Sampled Signal)

$$\begin{aligned} F(f^*(t)) &= F^*(\omega) = F\left[f(t) \sum_{n=-\infty}^{\infty} C_n \exp(jn\omega_s t)\right] \\ &= \sum_{n=-\infty}^{\infty} C_n F(\exp(jn\omega_s t) f(t)) \end{aligned}$$

but $\exp(jn\omega_s t) f(t) \xrightarrow{F} F(j\omega - jn\omega_s)$

$$\therefore F^*(\omega) = \sum_{n=-\infty}^{\infty} C_n F(j\omega - jn\omega_s)$$



From Spectrum of discrete (Sampled Signal):

⑥

① if $\omega_s > 2\omega_c$

we LPF \rightarrow Restore $f(t)$ ✓

② if $\omega_s = 2\omega_c$

Critical case for restoring $f(t)$

③ if $\omega_s < 2\omega_c$

we can't Reconstruct the Signal

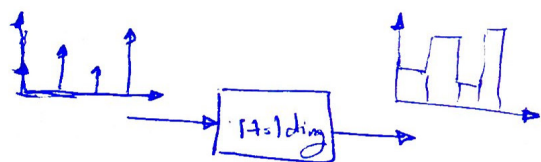
Shannon's Theory:

For right sampling Process ω_s (sampling frequency) must be equal or larger than twice ω_c

$$\omega_s \geq 2\omega_c$$

In Practical ω_s is chosen to be $(5-10) \cdot \omega_c$

* Holding Process



$$f_k(t) = f(kT) + f'(kT)(t-kT) + f''(kT) \frac{(t-kT)^2}{2!} + \dots$$

Taylor Expansion

where

- $f_k(t)$: The function expression in between (kT) and $(kT+T)$

- $f(kT) = f(t) |_{t=kT}$, $f'(kT) = \frac{d}{dt} f(t) |_{t=kT}$

\rightarrow If we choose only first term $\rightarrow f_k(t) = f(kT)$: Zero Order Hold



$$\therefore G_{ZOH}(s) = \frac{O/P(s)}{I/P(s)} = \frac{\frac{1}{s} - e^{-Ts} \cdot \frac{1}{s}}{1} = \boxed{\frac{1 - e^{-Ts}}{s}}$$